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COMMUNICATION ASPECTS OF PARALLEL PROCESSING

Cüneyt Özveren *

Laboratory for Information Decision Systems

Massachusetts Institute of Technology

Cambridge, MA 02139

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Abstract

Parallel processing was motivated by the need to solve very large computational problems, such as the numerical solutions of partial differential equations in the context of computational fluid dynamics, structural mechanics, image processing, etc. This report surveys recent literature on parallel processing algorithms for mainly rings, meshes, and hypercubes. These algorithms include vector, matrix computations, fixed point iterations and linear equation solvers. A group property of above topologies has also been explored in an attempt to develop tools for algorithms and performance analysis. Some special sparsity structure of the iteration dependencies has also been examined. A necessary and sufficient condition for the reducibility of a dependency matrix with sparse, nonzero extended diagonals has been derived.

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Parallel processing was motivated by the need to solve very large computational problems, such as the numerical solutions of partial differential equations in the context of computational fluid dynamics, structural mechanics, image processing, etc. This report surveys recent literature on parallel processing algorithms for mainly rings, meshes, and hypercubes. These algorithms include vector, matrix computations, fixed point iterations and linear equation solvers. A group property of above topologies has also been explored in an attempt to develop tools for algorithms and performance analysis. Some special sparsity structure of the iteration dependencies has also been examined. A necessary and sufficient condition for the reducibility of a dependency matrix with sparse, nonzero extended diagonals has been derived.

1 Introduction

This report surveys some recent literature on communication aspects of parallel processing. Parallel processing was born out of the need to solve very large computational problems, such as the numerical solutions of partial differential equations in the context of computational fluid dynamics, structural mechanics, image processing, etc. [1]. Suppose that we have a large scale computational problem and n identical computation units, or processors, connected via some communication network or "topology". It will be assumed that there is a relatively small number of powerful processors (like the Intel hypercube, 128 microprocessors [7]), as opposed to many simple processors (like the Connection Machine, 65000 processors). The processors cooperate in solving the computational problem. Each works on a different part of the problem and exchange information according to need or some schedule. The primary goal here is to minimize the total time needed to solve the problem.

1.1 Motivation

As a motivation, let us consider the following classes of problems:

Problem 1.1 Consider a fixed point problem given by the iteration:

$$x[k+1] = f(x[k]) \quad (1)$$

where $x[k]$ is an N dimensional vector. Given an initial value $x[0]$, the serial algorithm iterates until $\sqrt{(x[k] - x[k-1])'(x[k] - x[k-1])} < \epsilon$ for some given error tolerance ϵ . Suppose that A is a "dependency matrix", [1], for f in iteration (1), i.e. a_{ij} is nonzero if and only if f_i is a function of x_j . Then from a parallelization

point of view, the above problem is equivalent to the iteration:

$$x[k+1] = Ax[k] \quad (2)$$

A special case of Equation 1 is when x corresponds to the lexicographic ordering of variables defined on a grid where the iteration for each variable depends only on its immediate neighbors. A typical example of this is solving a partial differential equation, of say two variables, numerically by discretizing it over a two dimensional array of points and using a Jacobi type of an iteration, [14]. ••

Problem 1.2 Consider solving a set of N linear equations of N variables:

$$Ax = b \quad (3)$$

given b and a square, invertible (for simplicity) A . A numerically sound way to solve this is by first finding a QR decomposition of A and then doing a backsubstitution to find x (see [4]). QR decomposition reduces A to an upper triangular matrix using orthogonal transformations. These are simultaneously applied to b . The modified set of equations can now be solved iteratively by first calculating x_N and then using this to calculate x_{N-1} , etc. However, this approach may be undesirable for large and sparse A since it does not make use of any sparsity structure of A . An alternative approach is to use the conjugate gradient algorithm (see [4]). This algorithm is an iterative one that is guaranteed to terminate in N steps and executes the following in its main body:

$$\beta_k = r'_{k-1}r_{k-1}/r'_{k-2}r_{k-2}$$

$$p_k = r_{k-1} + \beta_k p_{k-1}$$

$$\alpha_k = r'_{k-1}r_{k-1}/p'_k A p_k$$

$$x_k = x_{k-1} + \alpha_k p_k$$

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$$r_k = r_{k-1} - \alpha_k A p_k \quad (8)$$

where $x_0 = 0, r_0 = b, \beta_1 = 0, p_1 = b$. Also, the algorithm terminates if $\sqrt{r[k]^T r[k]} < \epsilon$. Parallelization of this algorithm is considered in [7] in the framework of solving elliptic and hyperbolic equations using multigrid methods. They also consider a parallel preconditioner for the above algorithm, which is based on the Fast Fourier Transform. This preconditioner estimates the inverse of A above and uses it to improve the convergence rate of the conjugate gradient algorithm. ••

Suppose that for each problem above, the variables in question are distributed among some processors. Based on the following types of interconnections between these processors, or networks:

- Ring
- Mesh
- Hypercube

which are explained in Section 2.1, this report analyzes the following *standard computations*:

- Exchange information with another processor.
- Inner product of two vectors.
- Matrix-Vector multiplication.
- Matrix-Matrix multiplication.
- Matrix transpose.

These in turn motivate consideration of the following *standard traffic distributions*:

- One to one: This is mainly due to one particular processor exchanging information with another processor.
- Broadcast (Accumulation): In this case, one processor sends the same message to all the other processors. This comes up, for example, when the termination of an algorithm, decided on by a designated processor, is communicated to all the other processors. Accumulation comes up in calculating the inner product of two vectors such that their corresponding entries are stored in the same processor.
- Multinode Broadcast (Accumulation): In this case, all processors broadcast. For example, when a copy of a vector needs to be stored in all the processors, each processor broadcasts the entries it stores. Multinode accumulation is identical to multinode broadcast.
- Scatter (Gather): In this case, a particular processor sends a different message to every other processor. For example, a designated processor distributes a vector over all the other processors. Gather is the dual of scatter, in that a particular processor receives different messages from every other processor.
- Multinode Scatter (Gather): This is equivalent to matrix transpose if a number of rows (or columns) of a matrix are stored in each processor.

1.2 Overview and Outline

Saad and Schultz [12,13] present algorithms for the standard traffic distributions above with no specific application in mind. Bertsekas and Tsitsiklis [1] improve these algorithms and discuss applications to the standard computations above and

to numerical computations such as the fixed point problem. They also discuss implementation aspects and timing trade-offs very thoroughly. Both of these studies consider various classes of networks. McBryan and Van de Velde [7] concentrate on hypercubes and discuss general purpose algorithms, such as the standard computations above, and numerical solutions of elliptic and hyperbolic equations. They present results and performance plots for the various algorithms that they implemented on the Caltech and Intel hypercubes.

Unfortunately, there seems to be a lack of mathematical tools that could ease the analysis of different algorithms by taking advantage of common features of different classes of networks and the repetitious nature of the standard traffic distributions. In particular, it is not clear why certain networks are easier to analyze than others. Therefore, all the algorithms are essentially regenerated for different classes of networks and for different traffic distributions. I will try to improve on this by using group theory in describing certain classes of networks.

Section 2 defines the classes of networks considered in this report, presents best known algorithms for the standard traffic distributions and concludes by using group theory to derive some general results. Section 3 presents algorithms for the standard computations and the two problems discussed above. It also presents algorithms for iterative problems where the A matrix has some "regular" sparsity structure. Section 4 summarizes conclusions of the author and discusses possible extensions of the work surveyed in this report.

2 Results on Standard Traffic Distributions

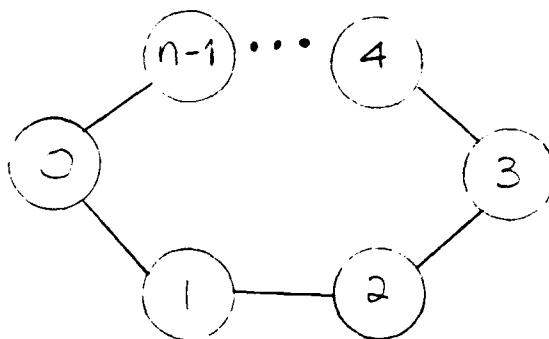
Efficient algorithms for standard operations motivated in the previous section are known for various classes of networks. A network is in general represented by a graph $G = (N, A)$ as a collection of nodes, N , and bidirectional arcs, A . An arc $(i, j) \in A$ is assumed to represent a full-duplex link (communication can proceed in either direction simultaneously) between the nodes $i, j \in N$. Also, I will assume that each link provides an error free message pipe (i.e. it is assumed that if an error occurs, it is detected and corrected in a negligible amount of time. This is *not* a very realistic assumption in general. However it is consistent with the assumptions of the papers surveyed and the consequences will be briefly discussed in Section 4). Note that the arcs constitute a measure of distance for the set of nodes, i.e. $d(i, j) = 1$ if $(i, j) \in A$ and in general the length of the shortest path between i and j . A path is specified as a sequence of neighboring nodes (distance 1 apart).

Suppose that a traffic distribution is specified in terms of messages of length $l_{i,j}$ to be sent from node i to node j for various pairs of nodes, that are not necessarily neighbors. In a communication network, a major problem, given a traffic distribution, is how to schedule these messages such that total communication time is minimized. This is termed *routing*. Routing is typically solved locally (at each node) since otherwise communication overhead to gather traffic information and then distribute routing results would be very excessive. This in general is a hard problem. For parallel processing algorithms considered in this report, traffic distribution is known a priori and it is assumed that the network belongs to a class which is "regular" in some sense. This enables one to precisely specify the path followed by each message and makes the timing analysis much easier. Different types of networks will be compared using the following criteria (as given in [1]):

- **Diameter:** Maximum shortest path distance between any pair of nodes. The time to transmit a message between any two nodes is at most a multiple (determined by the communication cost) of the diameter, assuming no queueing delays at the intermediate nodes.
- **Connectivity:** Provides a measure for the number of "alternate" paths connecting a pair of nodes. Node (arc) connectivity is defined as the minimum number of nodes (arcs) that must be deleted before the network becomes disconnected. If a network has arc connectivity k , then there are at least k parallel paths, i.e. paths with no arcs in common, between any pair of nodes (max flow - min cut theorem [8]). This is important for reliability purposes. Also, the communication between any pair of nodes can be parallelized by splitting the message into several parts and sending each on a different path. If there is no other traffic crossing these paths and the overhead cost for communication is low (not a valid assumption in general) then the total communication time for this message is reduced by a factor of k .
- **Flexibility:** Ability to efficiently run a broad variety of algorithms. For example, an algorithm may be most suitable for a particular type of network. If this network can be imbedded in another network (in a way that preserves the neighborhood structure), but perhaps not vice versa, then the latter network is more flexible than the former since it can run algorithms for both.

2.1 Classes of Networks

Let us review the following classes of networks and their properties:

Figure 1: n Node Ring

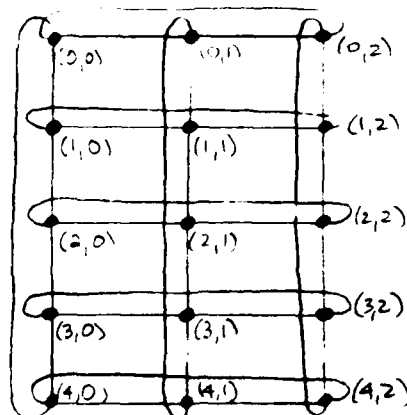
2.1.1 Rings

A ring with n nodes, is illustrated in Figure 1. It has diameter $n/2$, and arc connectivity 2. It will be assumed that the nodes are indexed consecutively by integers, as in the figure.

2.1.2 Mesh

An n node d dimensional mesh is a d dimensional array of $n = n_1 n_2 \cdots n_d$ nodes ($\{n_i\}$ is a positive, nonincreasing sequence of integers). It is assumed that the nodes are indexed as d -tuples (x_1, \dots, x_d) where each x_i is an integer modulo n_i . Only the immediate neighbors are connected, i.e. two nodes are connected only if their indices differ by one only in one coordinate. A wraparound mesh is one where two nodes are connected only if their indices differ by one modulo n_i only in some coordinate i . This report only considers wraparound meshes due to their symmetric structure. An example of such a mesh is illustrated in Figure 2. A wraparound mesh has diameter $(n_1 + \cdots + n_d)/2$ and arc connectivity $2d$.

Let us now consider imbedding a ring into a mesh with the same number of nodes. What is meant by imbedding is a one to one map that preserves neigh-

Figure 2: A Two Dimensional 5×3 Wraparound Mesh

neighborhood structure, i.e. neighbors are mapped to neighbors. For simplicity, let us consider two dimensional meshes. First recognize each, say, row of the mesh as a ring. Then, join the second row with the first to form another ring (for example, see Figure 3). To do this, break a link in the first row, and its counterpart in the second row. Join the nodes at the end points of these links. Next, join the third row with the second, in the same fashion. Note that here, a link different than the one picked in the previous case should be used. Proceed likewise with the other rings. At the third step, the link picked at the first step can be used, etc. Therefore, we never run out of links to pick. This can be extended to higher dimensions in a straightforward way. Also, using above approach, conditions can be derived for imbedding a ring in a mesh with more nodes.

2.1.3 Hypercube

Let H_d represent a d -dimensional hypercube, see for example Figure 4. It has 2^n nodes. Assume that these nodes are indexed by binary numbers of length d bits, B . Then two nodes are connected if and only if their indices differ by one bit. The distance between two nodes is given by the number of bits they differ in. This

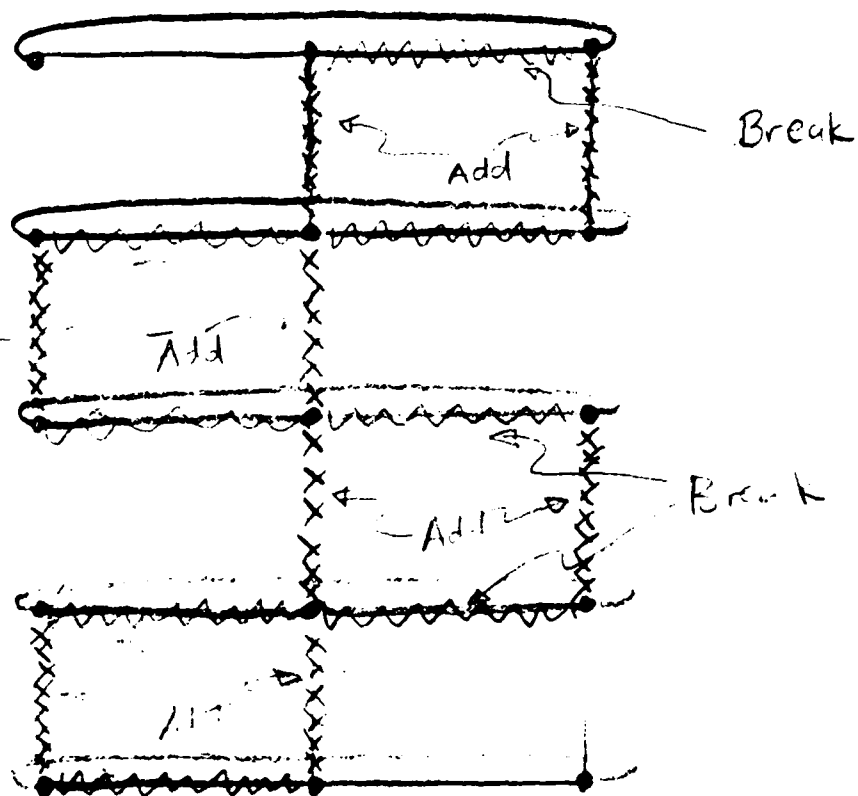
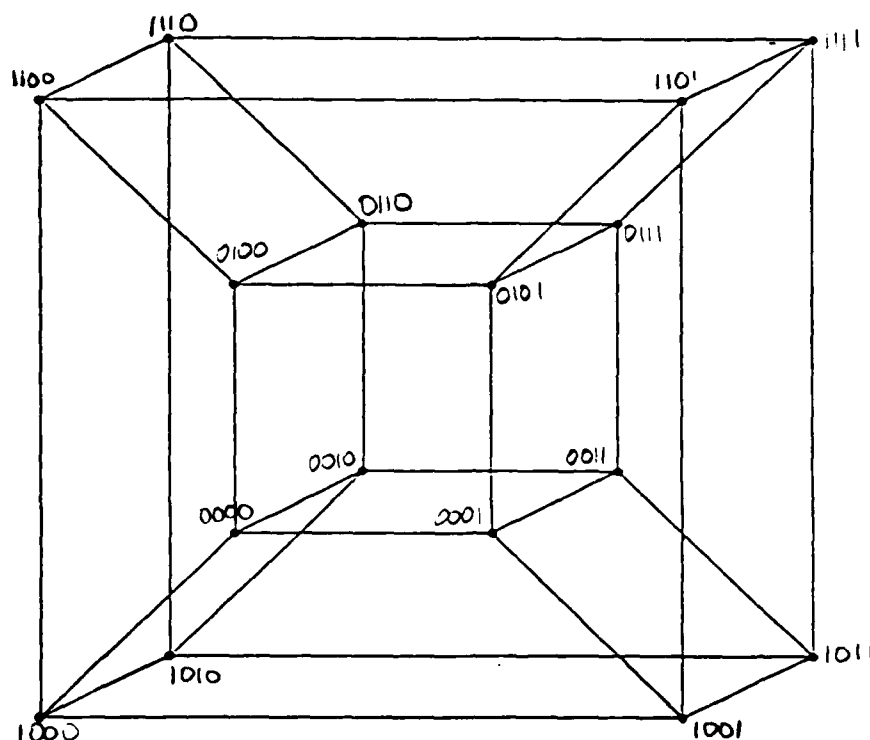


Figure 3: Imbedding a Ring in a 5×3 Mesh

Figure 4: Hypercube, $d = 4$

is termed the Hamming distance and it can be calculated using the exclusive OR function (XOR). XOR of two bits is 0 if they are equal and 1 if different. Let a and b be two nodes in a hypercube, then $a \text{ XOR } b$ is defined as the binary number obtained by XOR of corresponding entries of a and b . Hamming distance, $d_h(a, b)$, is the number of 1's in $a \text{ XOR } b$.

Any ring with an even number of nodes can be imbedded in a hypercube of equal or more nodes. First of all, note that a d dimensional hypercube consists of two $d-1$ dimensional hypercubes with corresponding nodes connected. Let us first consider rings with 2^d nodes. Clearly, we can perform this imbedding when $d = 2$. Suppose we can do it for some d . For $d + 1$ take two d dimensional hypercubes with a ring imbedded in each. Now, break the rings at some link in one hypercube

and the corresponding link in the other. Connect the corresponding nodes of the two hypercubes at either side of these links. We thus have an imbedding in the $d + 1$ dimensional hypercube (see Figure 5). This corresponds to traversing the nodes in one hypercube in the reverse direction of the other and generates the so called Binary Reflected Gray Codes [1,11,7]. In general, when the ring has an even number of nodes, r , it can be imbedded in a hypercube with equal or more nodes by joining the corresponding $r/2$ node portions of the rings at lower dimension. If a ring has an odd number of nodes then it *cannot* be imbedded in a hypercube since a hypercube has *no* cycles (i.e. no paths that finish at the nodes they start) with an odd number of nodes, [11]. A mesh with even n_i can be imbedded in a hypercube of dimension $\lceil \log n_1 \rceil + \dots + \lceil \log n_d \rceil$ (all logs are base 2). Using the imbeddings of a ring each column can be imbedded in a hypercube of dimension $\lceil \log n_1 \rceil$. Corresponding nodes of these hypercubes form a hypercube of dimension $\lceil \log n_2 \rceil$. Thus, they may be combined by n_2 node rings. Higher dimensions are imbedded similarly.

For all the network classes above I assume that communication can be carried out simultaneously on all links of a node in both directions. In [7], it is implicitly assumed that a node can transmit on only one link and receive on another at any given time. Although in practice it seems to be possible to transmit and receive on more than one link at a time, if not all links, I will consider both extremes for hypercubes. I refer to them as *type 1* and *type d*. Bertsekas and Tsitsiklis [1] consider both extremes for all networks.

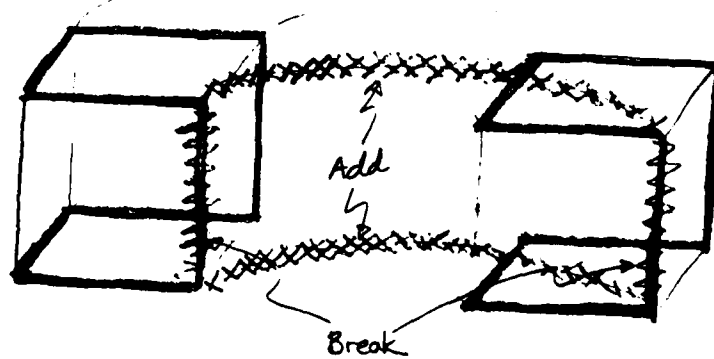


Figure 5: Imbedding a Ring in a Hypercube, $d = 4$

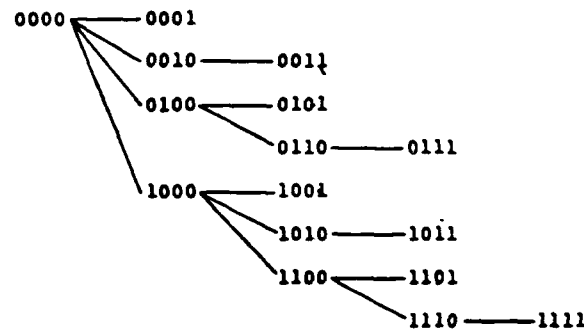


Figure 6: Tree Used for Broadcasting in a Hypercube of Dimension d

2.1.4 Trees

A tree is any connected network with no cycles. Note that a tree with n nodes has exactly $n - 1$ links (see for example Figure 6). Thus all trees have arc connectivity 1. The tree in Figure 6 has diameter 7. In general, trees are not very practical as physical networks due to their low connectivity. However, they are useful conceptually since they prevent data duplication in broadcast situations. Specifically, a spanning tree of a network, which is a subgraph of the network that includes all the nodes but has no cycles, is important (for example, Figure 6 is a spanning tree of a 4 dimensional hypercube).

2.1.5 Fully Connected

Examples of these are shared memory, crossbar switch (just like a phone system switch board), broadcast bus (every node is tapped on to a bus and only one can transmit at a time). I will not consider these networks since, as the number of nodes increases, shared memory and the switch network become impractical to implement, and time available for each node to transmit on a broadcast bus goes to zero. However, I should note that implementing switches on every node of a

hypercube to connect the incident links would relax the speed constraints on the local processor to keep up with transmissions on each link.

2.2 Standard Traffic Distributions

I will present algorithms for some standard traffic distributions on the network classes discussed above. I will assume that all messages are of unit length, and in most cases, present results in terms of the order of time needed in units of the transmission time of a unit length message. This will be denoted by $O(\bullet)$. If f is $O(g)$ but not $o(g)$ then f will be termed $O_s(g)$, i.e. strictly order g . If the best possible algorithm for a problem is known to be $O_s(g)$ and we have an algorithm that is $O(g)$ then that algorithm will be termed *efficient*. It is assumed that transmitting a message of length l over a link takes $c + lv$ units of time. Here c represents fixed costs, such as the propagation delay, and v represents variable costs, such as the transmission time.

2.2.1 One to one

In this case, one node sends a message to another node. The easiest way to do this for the rings is just to send the message along the shortest path, which requires time $(c + lv)s$ where s is the length of the shortest path. Note that this may be bounded by $(c + lv)n/2$. A better approach is to send part of the message through the alternate longer path, specifically a portion proportional to the length of the paths. In this case, total transmission time may be bounded by $(c + lv)n/4$. In [12], the message is divided into smaller messages and pipelined through the shortest path. This yields an equation for the total transmission time as a function of the number of messages, which can be minimized to yield the optimum number of messages.

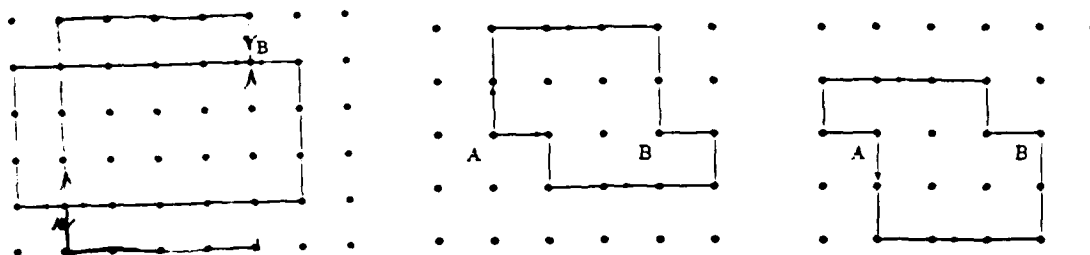


Figure 7: Parallel Paths on a Mesh

Yet another way is to send these smaller messages through both paths in a pipelined fashion. In general, one to one ring algorithms are $O_s(n)$.

For a mesh, it is shown in [12] that there are four parallel paths between any two nodes a, b of a two dimensional grid with wraparound (recall that we already know this fact from the Min-Cut Max Flow Theorem). In particular (see Figure 7),

- If a and b are aligned on the grid, then these paths can be chosen to be of length $d(a, b) + 2$ each.
- If a and b are aligned, in either the horizontal or vertical direction, but are not neighbors, then four paths can be chosen to be of length $d(a, b) + 4$.
- If a and b are neighbors, then four paths can be chosen to be of length $d(a, b) + 6 = 7$.

Thus, as in rings, the message could be divided into smaller messages, and can be sent through these parallel paths in a pipelined fashion. In general, one to one grid algorithms are $O_s(n_1)$.

For hypercubes, it is shown in [13] that there are d parallel paths between any two nodes a, b of a hypercube. If $d(a, b) = i$ then i of these paths are of length i and

the rest are of length $i + 2$. To illustrate this, note that a path can be chosen by equalizing the differing bits of a and b in some order. Suppose, this order is chosen in the direction from right to left. The first path is chosen starting from the first differing bit from the right and propagating to left. The second path is chosen by starting from the second differing bit, circulating and finally finishing with the first differing bit from the right. In this fashion, i parallel paths can be chosen. For the rest, first flip a bit that is identical in a and b . Then correct *all* the differing bits, there are $i + 1$ of them now, in any order but leave the previously correct bit to the last. Since there are $d - i$ bits common to both a and b , $d - i$ additional paths can be chosen with length $i + 2$ each. For type d , the message can be divided up and pipelined along these parallel paths. The resulting algorithm is $O_s(d)$. For type 1, pipelining is not useful but the message could still be divided up into pieces. Each piece should be sent to a different neighboring node one by one. It is slightly better if the nodes that are further away are started from. The algorithm is still $O_s(d)$ but the coefficient is doubled.

For all the above networks, pipelining is a final touch of optimization, otherwise they are of the same order.

2.2.2 Broadcast

In this case, a node sends the same message to *all* other links.

For rings, all nodes have two links and the best strategy would be to transmit on both sides of the broadcasting node. Here, the algorithm is $O_s(n)$. In [12], an algorithm of the same order of transmission time is given for the pipelined case.

For meshes, a similar strategy could be used. The message should be transmitted outward from the broadcasting node, but special attention should be paid to the

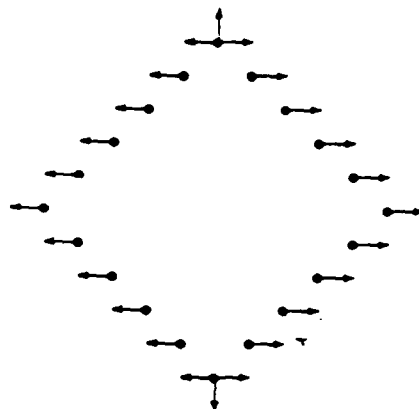


Figure 8: Broadcast Directions on a Mesh

direction of broadcast at each level so that no message is duplicated (see Figure 8 for illustration). Details of this algorithm and a variation based on splitting the message into a horizontal and vertical part are given in [12]. In both cases, the algorithm is lower bounded by the maximum n_i , i.e. it is $O_s(n_1)$.

For hypercubes, algorithms for this in [13,1,7] are all based on the same idea. There is no difference between type 1 and type d. The property that H_d consists of two H_{d-1} with corresponding nodes connected, is used. The broadcast node sends its message to the adjacent H_{d-1} . Now, we have two parallel broadcast problems on two H_{d-1} . Proceeding iteratively, the message is broadcast in d steps. An alternative representation of this algorithm is the tree in Figure 6. This tree is illustrated for node 0. The corresponding tree for any other node a can be generated by performing an XOR of a and each node on the tree. Note that this is a permutation of the tree such that the root is node a and the neighborhood structure is preserved. This is formally proved in Section 2.3.

The dual of the broadcast problem is accumulation, which can be executed in the same amount of time by reversing the steps of above algorithms.

2.2.3 Multinode Broadcast

In this case, all nodes broadcast their own messages.

For rings, the algorithm in [12] uses broadcasting in a round robin fashion around the ring. Each node receives a message from one direction and transmits it at the next step on the other direction. This requires $O_s(n)$ time, but additional savings, by a factor of 2 can be made by broadcasting in both directions simultaneously. Note that this is efficient since every node can receive two messages at every step and there is a total of n messages.

For meshes, the best algorithm in [12] is based on two passes. at the first step, a multinode broadcast is performed on each, say, horizontal ring of the mesh. Note that after the first pass, each node has n_1 messages corresponding to its horizontal counterparts. These messages are then broadcast along vertical rings, requiring a total of $O_s(n)$ time. Note that this algorithm is efficient since each node can receive 4 messages at a time and there is a total of n messages.

For hypercubes of type 1, multinode broadcast can be done using the vector shift algorithm of [7]. A ring with 2^n nodes is imbedded in a d dimensional hypercube. Multinode broadcast can be equivalently formulated as storing a vector of length n , whose elements are originally assigned to the processors in a one to one fashion, in each processor. The elements of this vector are shifted one by one along the ring until all processors receive the vector elements. This algorithm is $O_s(n)$ and efficient since each node can receive at most one message at a time and there is a total of n messages. For type d , there are various multinode broadcast algorithms in [13] and only one of them is efficient (Optimal Total Exchange Algorithm). It is $O_s(n/d)$, or $O_s(n/\log n)$, and efficient since there are d links from each node and a total of n messages to be received. However, this algorithm requires all nodes, at

each step, to scan the messages they currently hold to decide on which messages to send to their neighbors at that step. This requires processing time by the host processor and, depending on the particular architecture, it may take a considerable amount of time. A better algorithm is presented in Section 2.3.

2.2.4 Scatter, Gather

In the case of gather, one node receives a *different* message from every other node. Note that gather is a special case of multinode broadcast. Furthermore, every node has to receive n different messages. Thus, the algorithm for each network has a lower bound of n divided by the number of links from each node in that network. The multinode broadcast algorithms achieve this bound. Therefore, they are also efficient for gather ([13] has a gather algorithm for hypercubes which is not even efficient). Note also that a dual of gather is scatter. Efficient algorithms for scatter can be achieved by reversing the traffic streams of a gather algorithm.

2.2.5 Multinode Scatter or Matrix Transpose

In this case, every node sends a different message to every other node. For rings if we let each node scatter in turn, we have an $O(n^2)$ algorithm. Note that the messages corresponding to one node traverse $n/2$ links on the average. Since there are n nodes with n messages each and a total of n links, each link transmits an average of $n^2/2$ messages. Therefore, the above algorithm is efficient.

For meshes (the algorithm in [12] is wrong since for example, a processor on the diagonal never sends any messages to the processors in its row), say of dimension 2, first at each row we have a multinode scatter of messages destined for each column (in parallel), then we have a multinode scatter of these messages at each column (in

parallel). Thus we have a multinode scatter of messages n_1 units long over a ring of n_2 processors and a multinode scatter of messages n_2 units long over a ring of n_1 processors. Therefore, the algorithm is $O(n_1n_2^2 + n_2n_1^2) = O(n_1n)$. Note that the messages corresponding one node traverse $O_s(n_1 + n_2)$ links on the average. Since there are n nodes with n messages each and a total of $2n$ links, the above algorithm is efficient. For higher dimensional meshes, the corresponding algorithm will have the same order.

For hypercubes of type 1, the algorithm in [7] uses a matrix transpose interpretation. Suppose that we have an $n \times n$ matrix whose rows are distributed over H_d ($n = 2^d$). First divide the matrix into four blocks and exchange off diagonal blocks. This corresponds to exchanging messages between corresponding nodes of two H_{d-1} . Then divide each block into four sub-blocks and apply the above step to each block (see Figure 9). Iterating until we have 1×1 sub-blocks completes the algorithm. Note that at step i of the algorithm, $2^{d-i}2^{i-1} = 2^{d-1}$ messages are exchanged between each pair of nodes. Since there are d steps, the algorithm is $O(d2^d) = O(dn)$. For multinode scatter, each node transmits $O_s(n)$ messages, where each message goes through $O_s(d)$ nodes on the average. Since each node can transmit one message at a time and there are n nodes, the above algorithm is efficient. For hypercubes of type d, consider the following algorithm in [1]: This algorithm has two parts. First, a multinode scatter is performed in each H_{d-1} and in parallel each node sends the data for the other hypercube to the corresponding node (2^{d-1} messages). Then, a multinode scatter is performed on each H_{d-1} for these messages. It can be shown inductively that this algorithm is $O_s(n)$. Note that this algorithm can also be generated from the above algorithm for type 1. While off diagonal blocks are exchanged, diagonal blocks can be transposed (this is

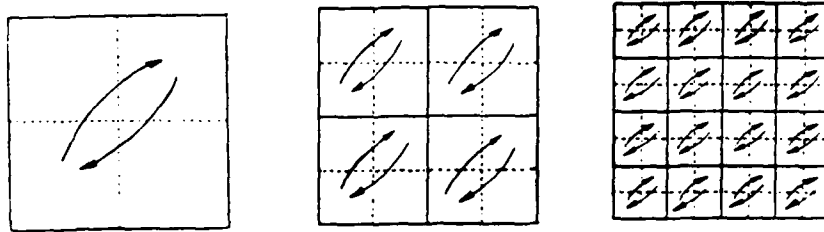


Figure 9: Matrix Transpose Algorithm for Hypercube Dimension 3, Type 1

	Ring	Mesh	Hypercube	
			Type 1	Type d
One to One	n	n_1	d	d
Broadcast	n	n_1	d	d
Multinode Broadcast	n	n	n	n/d
Scatter	n	n	n	n/d
Multinode Scatter	n^2	$n_1 n$	dn	n

Table 1: Orders of algorithms for standard operations on regular networks

the first part). Then, off-diagonal blocks are transposed (this is the second part). Note that this algorithm is also efficient.

The results presented in this section are summarized in Table 1.

2.3 Group Properties of Some Networks

It turns out that the nodes of rings, meshes and hypercubes can be indexed such that the set of indices is a group under some operation. Moreover, neighborhood structure and the shortest path between any pair of nodes can be described in a natural way using this operation. A group is defined as follows [10]:

Definition 2.1 A group is a set G with an associative binary operation $*$: $G \times G \rightarrow G$ such that:

1. There is an element e in G with

$$e * a = a \quad \text{for all} \quad a \in G$$

2. For every $a \in G$ there is a b in G with

$$b * a = e$$

(Then b is the inverse of a , denoted by a^{-1} .)

Moreover, a group is abelian if $*$ commutes. ••

For rings, let Z_n be the set of integers modulo n . Then Z_n is an abelian group under addition modulo n , call this $*_r$. The shortest path distance on a ring $d_r(a, b)$ for any $a, b \in Z_n$ is given by $\min(a *_r b^{-1}, a^{-1} *_r b)$.

For meshes, suppose that the nodes are indexed consecutively along each dimension by $Z_{n_1} \times \cdots \times Z_{n_d} = \mathcal{M}$. Let $a, b \in \mathcal{M}$ and $a = (a_1, \dots, a_d), b = (b_1, \dots, b_d)$. Then \mathcal{M} is an abelian group under addition, $c = a *_m b$, such that $c_i = a_i *_i b_i$ or $c_i = a_i + b_i \pmod{n_i}$. Then, the shortest path distance between any two points $a, b \in \mathcal{M}$ is:

$$d_m(a, b) = \sum_{i=1}^d \min(a_i *_i b_i^{-1}, a_i^{-1} *_i b_i)$$

For hypercubes, define a function $g_h : \mathcal{B} \rightarrow Z$ as the number of 1's in a binary number. Note that \mathcal{B} is a group under bit by bit XOR, $*_h$, and the shortest path distance, which is the same as the Hamming distance, is given by $d_h(a, b) = g_h(a *_h b)$. Note that the inverse of any element in \mathcal{B} is itself.

Consider a network such that the indices of nodes form an abelian group G under some operation $*$, such that the shortest path distance between any two

nodes $a, b \in G$ satisfies $d(c * a, c * b) = d(a, b)$ for all $c \in G$. Specifically, it is important that arcs map to arcs. Note that this is satisfied by rings, meshes and hypercubes. Suppose that we have an algorithm for broadcast from node e (the node whose index is the identity element in the group). In particular, this algorithm is specified as a set of distinct *directed* arcs A_i that transmit a message at step i of the algorithm. Let $(a, b) \in A_i$ and consider $c * (a, b) = (c * a, c * b)$. Since $d(c * a, c * b) = d(a, b) = 1$, $c * (a, b)$ is also an arc. Moreover, since broadcast from e sends the message from e to all other nodes and operating on these nodes by some node c is just a permutation of the nodes such that e maps to c , transmitting over the arcs $c * A_i$ at step i achieves broadcasting from node c . Therefore, it is sufficient to construct a broadcast algorithm from one node. The rest can be generated from this.

A straightforward approach to multinode broadcast is to let each node broadcast simultaneously using a broadcast algorithm. Recall that for the broadcast from node e , its message is transmitted over the arcs A_i at step i . This algorithm may be generalized to multinode broadcast such that at step i , the set of directed arcs $A_{i,a} = a * A_i$ ($*$ of each element of A_i by a) transmit the message corresponding to node a . Let (x, y) be an arc and $r_i(x, y)$ be the number of $A_{i,a}$ that contain (x, y) , i.e. $r_i(x, y)$ is the number of messages that need to be sent on arc (x, y) at step i . Thus step i takes

$$r_{\max,i} = \max_{(x,y) \in A_i, a \in G} r_i(x, y)$$

units of time. The sequence of sets A_i should be picked such that total completion time of the algorithm, $\sum_i r_{\max,i}$ is minimized. I claim the following:

Proposition 2.2 *Given A_i , let $(x, y) \in A_i$ and*

$$R_i(x, y) = \{(z, w) \text{ such that } (z, w) \in A_i \text{ and } x * y^{-1} = z * w^{-1}\}$$

Then,

$$r_i(x, y) = |R_i(x, y)|$$

Proof: First of all, let $(z, w) \in R_i(x, y)$ and $a = y * w^{-1}$ (note that $a \in G$). Then, $(a * z, a * w) = (y * w^{-1} * z, y) = (x, y)$ (using commutativity). On the other hand, Let $(x, y) \in A_{i,a}$ for some a . Then, $(x, y) = (a * z, a * w)$ for some, but only one, $(z, w) \in A_i$. We have, $x * y^{-1} = a * z * w^{-1} * a^{-1} = z * w^{-1}$. ••

Note that we can apply the above result to all $A_{i,a}$ individually. But also, $R(a * x, a * y) = a * R(x, y)$. Thus,

$$r_{\max,i} = \max_{(x,y) \in A_i} r_i(x, y)$$

i.e. we only need to look at the arcs for node e , and we have:

$$r_{\max,i} = \max_{(x,y) \in A_i} |R_i(x, y)| \quad (9)$$

In short, we wish minimize the number of arcs that carry traffic in the same direction at the same step. For hypercubes, these arcs correspond to arcs that flip the same bit. Note that the tree in Figure 6 would lead to a multinode broadcast in $O(n)$ steps. Conceptually, we should be able to generate a tree with about n/d levels and at most d arcs at each level such that all arcs at the same level flip a different bit. I propose the following: Consider a tree generated by d major steps and some minor steps in each major step. At major step i , all nodes with labels that have exactly i bits equal to 1 are connected to the other elements of the tree generated upto step i . In each minor step corresponding to step i , d of these nodes are connected to *any* previously generated element of the tree such that each arc corresponds to a different bit flip. The tree is generated when finally node 1^d (a

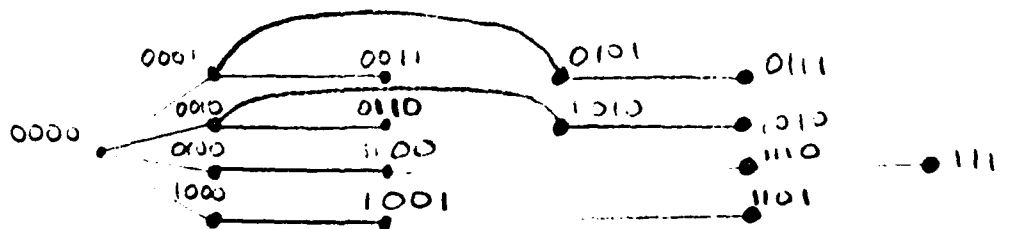


Figure 10: Multinode Broadcast Tree for Hypercube of dimension 4

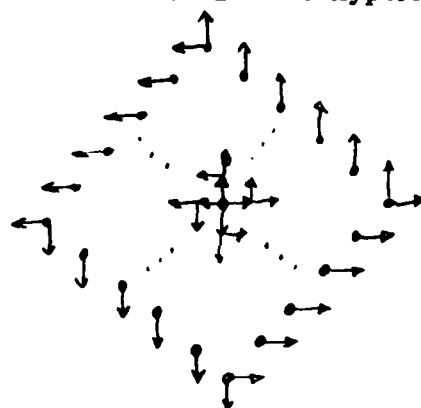


Figure 11: Multinode Broadcast Tree for a Mesh

node with index all ones) is connected. The total number of minor steps is bounded by $\binom{d}{i}/d + 1$ (see Appendix B). Such a tree for $d = 4$ is illustrated in Figure 10. This algorithm is $O_s(n/d)$ and efficient.

Note that above procedure is a proof that the multinode broadcast algorithm, described previously, for rings work. Similarly, a broadcast tree may be generated for a mesh, as in Figure 11, that produces an efficient multinode broadcast algorithm (compare to Figure 8). Note that this approach could also be applied to multinode scatter problems or any other problem such that the transmissions for all the nodes can be generated from the transmissions for one node using the group operation.

3 Applications

This section presents applications of standard traffic distributions of the previous section to vector, matrix computations, linear equation solvers and fixed point problems.

3.1 Vector Computations

Let us first consider computation of the inner product of two vectors. This is important for calculating the termination condition of a relaxation algorithm (for example checking to see if the norm of the error vector is small enough) and also constitutes a basis for matrix-vector multiplications. Suppose that two given vectors of length n each are distributed over n processors in a way that corresponding entries of each vector reside in the same processor. Let each processor calculate that portion of the inner product corresponding to the entries it holds. It then sends its result to another processor which adds it to its own computation and sends the result to another processor etc. The result is finally summed up at a designated processor. Thus, we have an accumulation (of the designated processor) problem. In relaxation algorithms, when this inner product is the norm of the error vector, if this norm is small enough then the designated processor sends a termination message to all the other processors. This is a broadcast problem.

Next let us consider a problem of shifting a vector by a certain amount. This is particularly interesting and important (since it constitutes a basis for matrix-vector multiplication, etc.) for type 1 hypercubes. In this case, ordering of distribution over the processors of the hypercube is important. In particular, assuming that the vector has n elements, each consecutive entry of the vector will be distributed

according the embedding of a ring into the hypercube (as generated using Binary Reflected Gray Codes). As a result of this imbedding, nodes that have logical distance 2^k on the ring, for some k , have physical distance 2 on the hypercube [1]. The algorithm in [7] uses this to construct a logical hierarchy of the ring in the hypercube (see Figure 12). The important issue here is that at level k of this hierarchy, there are 2^k *parallel* cycles corresponding to connections of each node with the nodes of logical distance 2^k . Thus, it takes two units of time to shift a vector by 2^k elements, for any $k > 1$ and one unit of time for $k = 1$. Consider shifting a vector arbitrarily by some amount $s > 0$. We can assume that $s < 2^d$ since shifting by 2^d is the same as not shifting at all. Also, we can assume that $s \leq 2^{d-1}$ since shifting in one direction by s is the same as shifting in the other direction by $2^d - s$ and when $s > 2^d$ we could save on the number of shifts by shifting in the other direction. Let us take the binary expansion of s , i.e. let $s = b_{d-2} \cdots b_0$. In [7], they achieve this shift by a combination of shifts depending on which bits of s are set. Since shifting by one requires one step and the rest require two steps each, they have an algorithm that works in $O(2(d-2) + 1)$ time in the worst case. Note that for a shift by 7, this requires shifting by 4, shifting by 2 and finally shifting by 1. However, it would take less time if this vector was first shifted by 8 and then shifted in the reverse direction by 1. To generalize this, one needs to solve a shortest path problem. It can be shown that any shift can be done in at most d steps, an improvement by a factor of two (see Appendix A).

3.2 Matrix Computations

Let us first consider how to store a matrix in n processors. Let us assume that the matrix is $n \times n$. There are three main forms of storage considered in [7] and [1]:

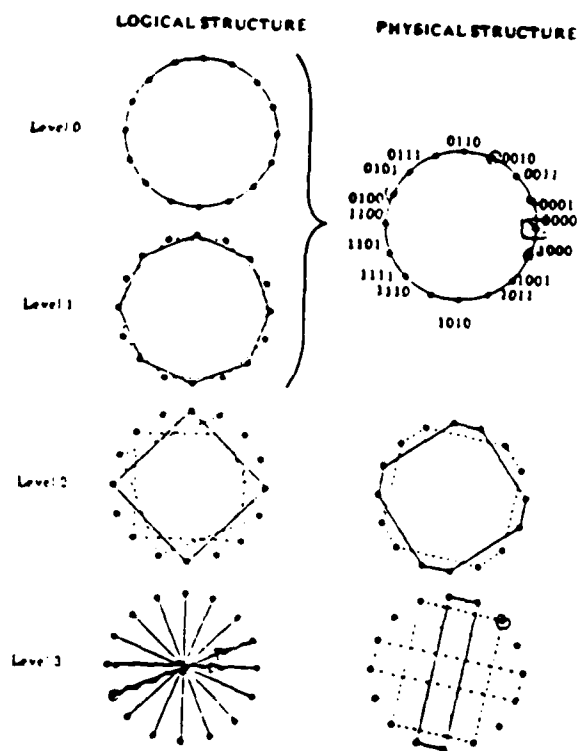


Figure 12: The logical hierarchies of rings in a 16-node hypercube and the communication channels used to implement them.

- By rows: Store each row in a processor.
- By columns: Store each column in a processor.
- By diagonals: The matrix A is converted into diagonal form D (see Figure 13) and D is stored by columns.

Note that conversion from storage by rows to by columns and vice versa requires multinode scatter. For each case above, we have the following algorithms:

- Suppose that the matrix is stored by rows and each processor has a copy of the vector. Then in $O(n)$ computation time the product is calculated and entry p of the resulting vector is stored in processor p . If we are running a relaxation algorithm, we need to redistribute the entries of the vector to all the processors. Thus we do a multinode broadcast.
- Suppose that the matrix is stored by columns and processor p has entry p of the vector. Each vector multiplies the column and the vector entry it stores. Then in $O(n)$ computation time, we have n vectors that are distributed over processors and need to be summed up. Since we want processor p to have entry p of the resulting vector, each processor needs to gather corresponding entry from every other processor. Here we have multinode gather. Note that alternatively, the entry may be added up at the intermediate processors. In this case, we have a multinode accumulation which would make the total cost of running the algorithm same as the storage by rows. However, as each processor accumulates its message, intermediate nodes need to add the different messages destined for the same node and form a new message. Thus, in practice, depending on the particular architecture this form of storage could

$$A = \begin{bmatrix} 00 & 01 & 02 & 03 & 04 & 05 \\ 10 & 11 & 12 & 13 & 14 & 15 \\ 20 & 21 & 22 & 23 & 24 & 25 \\ 30 & 31 & 32 & 33 & 34 & 35 \\ 40 & 41 & 42 & 43 & 44 & 45 \\ 50 & 51 & 52 & 53 & 54 & 55 \end{bmatrix} \quad D = \begin{bmatrix} 00 & 11 & 22 & 33 & 44 & 55 \\ 50 & 01 & 12 & 23 & 34 & 45 \\ 40 & 51 & 02 & 13 & 24 & 35 \\ 30 & 41 & 52 & 03 & 14 & 25 \\ 20 & 31 & 42 & 53 & 04 & 15 \\ 10 & 21 & 32 & 43 & 54 & 05 \end{bmatrix}$$

Figure 13: Conversion of A to Diagonal Form D

result in longer running time than the previous one, but perhaps the same order of magnitude.

- Suppose that the matrix is stored by diagonals and processor p has entry p of the vector. Then an algorithm can be constructed that consists of n successive shifts by one, and $O(n)$ computation before each shift (see [7]). They argue in [7] that this storage is portable in the sense that it does not bias A towards either row or column storage. This seems to be a reasonable argument for hypercubes of type 1. However, if the same storage is used for hypercubes of type d , corresponding calculation would take time by a factor of d larger than that of either row or column storage.

When multiplying two matrices A and B , the most favourable case is when A is stored by rows and B is stored by columns. The algorithm would be to perform n successive matrix-vector multiplications. The resulting matrix is stored by rows, in a total of $O(n^2)$ computation time and n multinode broadcasts.

3.3 Fixed Point Problems with Sparse A

Fixed point problems can simply be solved by a repeated application of the matrix-vector product considered above. However, for sparse A , this approach may not be very efficient. Some forms of sparsity structure are examined below. When necessary, applications to hypercubes of type d are used for illustration. In what follows, it is assumed that A is a square matrix of dimension $N \times N$ where $N = mn$ and n is the number of processors as usual.

3.3.1 Banded

Suppose that A is distributed among the processors in blocks of m rows, and similarly, x is distributed in blocks of m elements. Moreover, corresponding blocks of A and x reside in the same processor (for example, first m rows of A and first m elements of x are stored in the same processor) and the consecutive blocks are stored in neighboring processors (for example, for the hypercube, the imbedding of a ring into the hypercube would be used). To define what is meant by *banded* let us first consider the notion (of [7]) of an *extended diagonal*. An extended diagonal p of A is the set of entries $\{a_{ij}$ such that $j - i = p$ or $i - j = n - p\}$. The symmetric counterpart of this diagonal is the set of entries $\{a_{ij}$ such that $i - j = p$ or $j - i = n - p\}$. When an extended diagonal is nonzero, it will be assumed that its symmetric counterpart is also nonzero. A matrix A will be termed banded with width b if extended diagonals $1, \dots, b$ and their symmetric counterparts are nonzero and the rest of the matrix is zero. It will be assumed that b is a multiple of m , say $b = rm$. Then the corresponding fixed point iteration can be computed by r consecutive shifts in both directions. Each shift consists of transmitting m elements of x to a neighbor. Since the links are bidirectional, shifts to both directions can be

done in parallel and thus the algorithm takes $O(mr)$ communication time. There is also $O(m)$ computation after each shift. Note that, on a hypercube of type d, this method will be preferable to the matrix-vector multiplication method of the previous section if roughly, $r < n/d$.

3.3.2 Mesh

In certain cases, such as the 5-point discretization of a partial differential equation using the Jacobi method (see [14]), the "dependency graph" (i.e. the graph generated by modelling each variable as a node, and defining an undirected arc between nodes i and j for all nonzero a_{ij} of the dependency matrix) may be a wraparound mesh. Distribution of variables among n processors is modelled, in [7], as a problem of dividing a rectangle into n regions with minimum perimeter to area ratio. They assume that the mesh is $N_1 \times N_2 = n_1 p_1 \times n_2 p_2$ and the number of available processors is $n_1 n_2$. In this case, they argue that, to minimize communication, the mesh should be divided into squares with $p_1 p_2$ points each, if possible (otherwise near square rectangles). Each processor is assigned one of these rectangles. The dependency graph for the processors is then a wraparound mesh itself. At each iteration of the fixed point computation, each processor exchanges data with its horizontal and vertical neighbors. Recall that wraparound mesh consists of rings in horizontal and vertical directions. Thus, a shift by one in each direction of all horizontal and vertical grids may be done in parallel. This algorithm takes roughly $O(\sqrt{p_1 p_2})$ communication time (on the appropriate mesh or type d hypercube).

In 9-point discretizations, the dependency graph includes diagonal connections (see Figure 14). Suppose that this graph is $N \times N$ and $n = p^2$ for some integer p that divides N . As before, the graph could be divided into $r \times r$ squares where $r = N/p$.

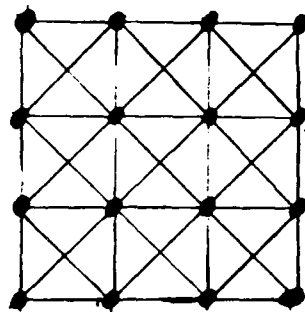
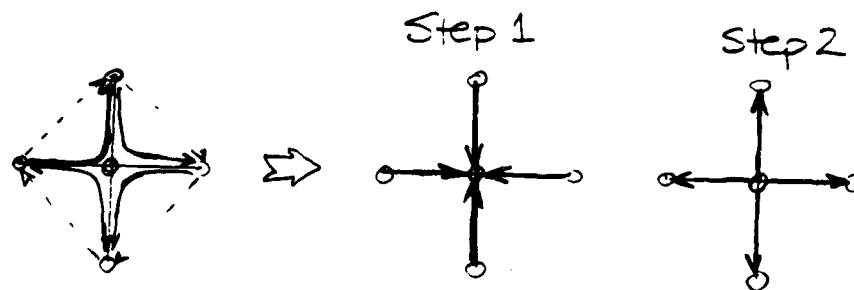
Figure 14: 4×4 Mesh Corresponding to a 9-Point Discretization

Figure 15: Decomposition of Diagonal Exchange

In this case, we have a similar $p \times p$ graph for which data corresponding to r points need to be exchanged on the vertical and horizontal directions, and 1 point on the diagonal directions, at each iteration. The data exchange can be performed on a $p \times p$ mesh of processors in two steps. Suppose that diagonal exchange is divided into its horizontal and vertical components as in Figure 15. At the first step, horizontal and vertical exchange takes place. This also covers the first components (horizontal or vertical) of the diagonal exchange. At the second step, the second components of diagonal exchange take place. Thus data exchange completes in $r + 1$ time units.

3.3.3 Scattered Off-Diagonals

This section analyzes fixed point iterations where A , $N \times N$, has p extended diagonals, r_0, r_1, \dots, r_{p-1} nonzero (see Figure 16). In [1] it is noted that a special case of the above where $r_i = (i+1)k$ and N is a multiple of k (see Figure 17), is undesirable when the matrix-vector multiplication method above is applied. However, this problem can be divided into k disjoint problems, each indexed by $i \in \{0, \dots, k-1\}$. Problem i corresponds to iteration of variables $\{i, k+i, \dots, N/k-1+i\}$. For the general case, group theory is useful. Let G be the group of integers modulo N with group operation of addition modulo N . In general, decomposition into disjoint problems relies on finding $\mathcal{R}_p = [r_0] * [r_1] * \dots * [r_{p-1}]$, where $[a]$ denotes the set of all powers of a , which is a subgroup of G , [10], and $[a] * [b] = \{a^i * b^j, \text{ for some } i \text{ and } j\}$. The disjoint problems are \mathcal{R}_p and its cosets, [10]. I first make the following claim:

Proposition 3.1 $\mathcal{R}_p = [r_0] * [r_1] * \dots * [r_{p-1}] = [q_p]$ where $q_p = \gcd(r_0, \dots, r_{p-1}, N)$.

Proof: I prove this inductively. First, $[r_0] = [q_1]$: Since $r_0 = q_1^s$ for some integer s , $[r_0] \subset [q_1]$. Let $a \in [q_1]$ then $a = q_1^h$ for some h . Note that $h \in G$. Since s , above, and N are relatively prime and $s \in G$, $[s] = G$ and thus for any $h \in G$, there exists some $w \in G$ such that $h = s^w$. Therefore, $a = q_1^{s^w} = r_0^w \in [r_0]$.

Suppose, $\mathcal{R}_p = [q_p]$, then $\mathcal{R}_{p+1} = [q_p] * [r_p] = [q_p] * [s]$ where $s = \gcd(r_p, N)$. On the other hand, $N = q_{p+1}(q_p/q_{p+1})(s/q_{p+1})w$ for some w . Thus,

$$\begin{aligned} \mathcal{R}_{p+1} &= [N/(s/q_{p+1})] * [N/w] * [N/(q_p/q_{p+1})] * [N/w] \\ &= [N/(s/q_{p+1})] * [N/(q_p/q_{p+1})] * [N/w] \\ &= [q_{p+1}] \end{aligned}$$

••

Note that there are exactly q_p cosets of $[q_p]$. To summarize, we have the following:

Proposition 3.2 *For a fixed point problem, or a matrix-vector multiplication problem, where A is structured as above, the problem can be divided into q_p disjoint problems (i.e. A is reducible to a block diagonal form) and moreover, the indices of the variables that belong to each problem are given by $[q_p]$ and its cosets.* ••

If the number of available processors, n , is less than or equal to q_p then one or more subproblems can be assigned to each processor appropriately. (If $n < q_p$, depending on the problem, it may be better to distribute some of the subproblems over all the processors to balance the computation load on the processors.) If $n = q_p$, then there is no communication cost, and the computation executes n times faster than the serial one. If $n > q_p$, say $n = mq_p$ for simplicity, then m processors could be assigned to each problem. In this case, if possible, either band or grid structure could be exploited as above or a straightforward matrix-vector multiplication type of an algorithm could be applied.

In general, comparison of a parallelized fixed point iteration to its serial version is hard because the performance is strongly coupled to the sparsity structure of A and how it is exploited. In the case of linear iterations with dense A , a serial algorithm takes $O(m^2n^2)$ computation steps. For hypercubes of type d , parallel version takes $O(mn/d)$ communication steps and $O(m^2)$ computation steps. Typically, computation is much faster than communication (for example for Intel hypercube [7]). Thus, for moderate m , communication will dominate computation and parallelization will be beneficial when unit communication cost to unit computation cost is roughly less than md/n . For large m , computation will dominate and parallelization will be $O(n^2)$ faster than the serial version. Another criterion given in [1] and

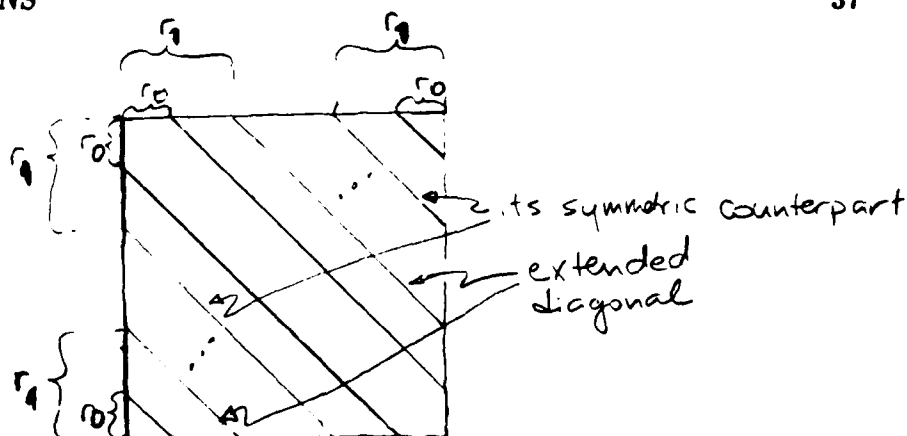


Figure 16: Scattered Extended Diagonals (entries nonzero only along the lines indicated)

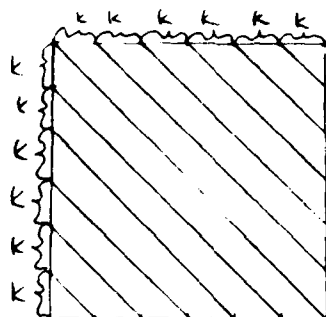


Figure 17: A Special Case of Scattered Extended Diagonals

[7] is the communication penalty, which is defined as the ratio of communication time per iteration to computation time per iteration. In this case, communication penalty is $O(n/dm)$ which approaches zero as m increases implying that processors are fully utilized.

3.4 Linear Equation Solvers

3.4.1 Orthogonalization

Let us first consider the QR decomposition (Q is not constructed explicitly, but the same transformations are applied to b). Suppose that A is $mn \times mn$ and stored by rows. In this case the dimension of the matrix being worked on decreases with each step. To preserve effective use of each processor, suppose rows $p, p+n, \dots, p+(m-1)n$ are stored at processor p . At each step, one column is orthogonalized. At step i , each processor first orthogonalizes that piece of the current column it stores, and this takes $O((m-i/n)(mn-i))$ computation time. Then the remaining part of the column (one entry per processor) need to be orthogonalized. However, the resulting transformation must to be applied to the rest of the rows. This can be achieved either using Modified Gram-Schmidt or a series of Givens transformations, [4], over a tree (such as the tree in Figure 6). In either case, the required communication time is $O(d(mn-i))$ and the computation time is also $O(d(mn-i))$. Summing this over mn steps, we get $O(dm^2n^2)$ communication time and $O(m^3n^2)$ computation time. Backsubstitution takes $O(mn)$ time. The serial version of this algorithm takes $O(m^3n^3)$ computation steps. Thus parallelization is favourable when unit communication time to computation time ratio is roughly less than mn/d . Communication penalty is $O(d/(m+n))$ which approaches zero as m increases.

3.4.2 Conjugate Gradient

In this section, I calculate the execution time of Equations (4-8) of the first section. All vectors and the matrix A , $mn \times mn$, are assumed to be distributed by rows. The calculations below assume a type d hypercube. Equation (4) relies on computing

a single inner product. Recall that this takes $O(mn/d)$ time. Equation (5) is immediate after Equation (4). Equation (6) requires a multinode broadcast (Note that $p'_k A p_k$ can be computed at the same time with $A p_k$ since every processor has a copy of p_k), and thus takes $O(mn/d)$ communication time. If A is sparse, this may of course be further improved, if possible, using the approach of Section 3.3. Equations (7) and (8) are immediate after Equation (6). Thus each iteration takes a total of $O(mn/d)$ communication time and $O(m^2 n)$ computation time. In the serial version each iteration takes $O(m^2 n^2)$ computation steps. Therefore, if the communication to computation ratio is roughly less than mnd , it will be preferable to parallelize. Note that the iterative version is faster, by a factor of $O(d^2)$, than the QR decomposition approach above even in the worst case convergence of mn steps. Furthermore, the iterative version can be made faster if A is sparse. Note also that the communication penalty is $O(1/md)$ and approaches zero as m increases.

4 Conclusions and Further Questions

This report has surveyed part of the recent literature on the communication aspects of parallel processing. Among the topologies considered, hypercubes have the best properties in terms of diameter, connectivity, and flexibility. For the standard operations, efficient algorithms are known for the popular topologies. From a practical point of view, matrix vector product seems to be the most important computation. This computation, for dense A , is rather straightforward. However, exploiting the sparsity structure of A is a hard problem in general.

As a tool for exploring structures of "regular topologies" and analysing timing properties of algorithms for these topologies, the application of group theory was proposed and carried out to some extent. An interesting result was presented on the reduceability of a sparse matrix with nonzero extended diagonals. One could argue that such matrices may not be found in practice since if the matrix is reducible, then this perhaps can be recognized at the modelling stage. However, the matrix may be "almost reducible" in the sense that appropriate extended diagonals are $O(\epsilon)$ for some small ϵ . This would correspond to weakly coupled subproblems of the large scale problem. Then a processor can be assigned to each subproblem and a two time scale update could be carried out, i.e. variables corresponding to each subproblem are updated frequently with respect to other variables in that subproblem and are updated at a slower rate (perhaps bounded by communication time) with respect to the variables in other subproblems.

The algorithms considered in this report were essentially synchronous. Although the processors operate asynchronously in principle, each processor waits for all the variables necessary for an update before performing that update (local synchronization, [1]). This may potentially lead to some reliability problems. In particular, if

a message is lost while being transmitted (for example when a link goes down), depending on how these losses are treated by the particular network, it may be delayed arbitrarily. This would in turn lead to an arbitrary delay in the update of the particular processor the message is destined for. Since other processors will wait for the results of this update in their next update, execution of the algorithm would be delayed arbitrarily. A potentially more robust approach is the use of asynchronous algorithms, [1]. Typically, in this case, each processor performs the updates as many times as possible regardless of whether it has the most recent values of the variables in other processors. A major drawback of these algorithms is that their convergence properties are much harder to analyze.

Another aspect of reliability is the extent of damage caused by a failure, such as a link or node crash, on the overall communication properties of the network. For example, when a link crashes, a natural remedy would be to reroute the packet through an alternate path. However, there could conceivably be other packets using the links on this path, and they will be delayed due to the rerouted packet. For example, for the algorithms considered in this report, the precise scheduling of the messages would be disrupted and communication time estimates would not be valid anymore. Note that the nodes are generally aware of the current traffic distribution. The network may use this to its advantage by implementing a different (perhaps pre-calculated) schedule.

I believe that the recent developments on the control of Discrete Event Dynamic Systems, [2,3,5,6,9], may be used to address robustness problems. In particular, the work of Ramadge and Wonham [9], Lin and Wonham [6] and Cieslak et al. [2] could be helpful in designing a local control that reacts to link failures. The work of Cohen et al. [3] could be used in analyzing the effects of the traffic distribution and

the times at which the messages are introduced into the network, on the queue sizes. Finally, perturbation analysis of Ho [5] might be useful in analyzing the effects of small changes in the scheduling on the overall performance.

A Minimum Distance on a Hierarchical Ring

This appendix formulates the problem of finding the optimal combination of shifts on a hierarchical ring in a hypercube to achieve an arbitrary shift, as a shortest path problem and shows that this can be done in at most d steps. Consider the following scalar linear system:

$$x[k+1] = 2x[k] + u[k]$$

where $u[k]$ is restricted to 1 (forward shift), 0 (no shift), -1 (reverse shift). Let x be the desired amount of shift and consider a minimum energy control problem to reach x in the above system in some number of steps. Let us plot the state space of this system and formulate this as a shortest path problem where absolute values of $u[i]$ are the arc lengths. Figure 18 illustrates this for a 32 node ring. Note that initial input of -1 is omitted due to the symmetric nature of the state space and we do not need to consider any states larger than 16 since we can equivalently achieve it by reversing the shifts. The numbers in squares beside the states are the shortest path lengths. For example, a shift of 7 can be achieved by a shortest path through 0, 1, 2, 4, 7, with a corresponding length of 2, and input string 1, 0, 0, -1 which corresponds to a forward shift by $2^3 = 8$ and a reverse shift by 1. The graph for larger rings can be generated by building on this ring. A ring with 2^d nodes will have a graph with $d - 1$ levels. At each level, maximum shortest path length of an even numbered state is the same as maximum shortest path length of an odd numbered state in the previous level. On the other hand, maximum shortest path length of an odd numbered state is one plus maximum shortest path length of an even numbered state in the previous level. It can be shown inductively that at level k , maximum shortest path for an even state is $k/2$ and odd state is $k/2 + 1$ if k

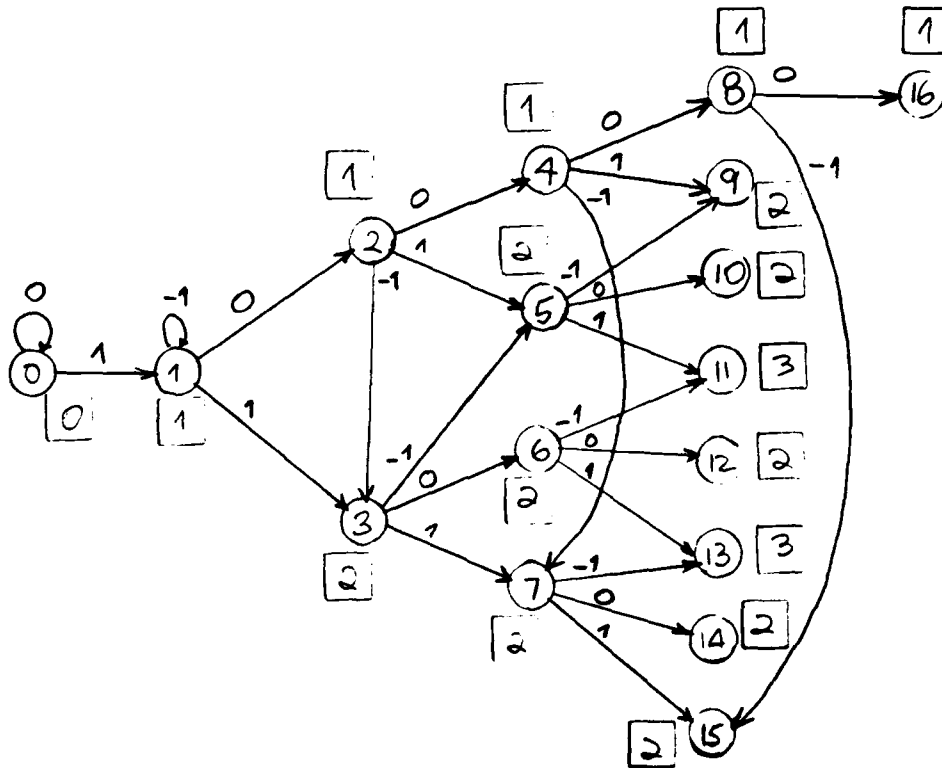


Figure 18: State Space for a 32 Node Ring

is even, and $(k + 1)/2$ for all states if k is odd. Recall that shifts that are nonzero powers of 2 take two steps. Also, note that exactly one shift of an odd numbered node will be a single step shift and all shifts of even numbered nodes take two steps. Therefore, it follows that arbitrary shifts require at most d steps for hierarchical rings with 2^d nodes on a hypercube.

B Multinode Broadcast Tree for Hypercubes

To prove the existence and the timing properties of a multinode broadcast tree for hypercubes as derived in Section 2.3, I will use the equivalence classes defined in [13]. Let $A(d, i)$ be the set of d bit binary numbers with exactly i ones. Divide this

set into disjoint equivalence classes using the relation: $a \mathcal{S} b$ if and only if $a = S^j b$ for some positive integer j , where S is the left cyclic shift operator.

In each minor step of the tree construction, an equivalence class is connected. Note that in all the equivalence classes, there is an element with a one in the first bit and other elements are shifts of this element. Given an element of $A(d, i)$, if we change a one entry of this number to a zero, the resulting number is an element of $A(d, i-1)$. Therefore, all elements of an equivalence class of $A(d, i)$ can be connected to elements of $A(d, i-1)$ in a way that each connection flips a different bit. To prove that the number of minor steps in a major step is bounded by $\binom{d}{i}/d+1$, it suffices to prove that at most one equivalence class of $A(d, i)$ has less than d elements (others have exactly d elements).

Proposition B.1 *There are exactly $\lceil \binom{d}{i} \rceil$ equivalence classes of $A(d, i)$.*

Proof: Let us first prove that there is at most one equivalence class with less than d elements in it. Let $a \in A(d, i)$ and j be the smallest integer such that $a = S^j a$. Suppose $j < d$ and without loss of generality, assume that a has a one as the rightmost bit. Denote the positions of ones in a by integers modulo d such that e is the position of the rightmost bit. By the property of cyclic shift, ones are located at $[j]$, that is, at $j, 2j, \dots$ etc. By the proof of Proposition 3.1, $[j] = [p]$ where $p = \gcd(n, j)$. Also, $p^i = e$ and thus $pi = d$. Therefore, given i , p is unique and furthermore, it exists if and only if i divides d .

Using the fact that $A(d, i)$ has $\binom{d}{i}$ elements, we achieve the desired result. $\bullet\bullet$

Note that the tree in Figure 10 can be improved by filling the gaps (see Figure 19). However, if d is prime, then all equivalence classes (except for the one corresponding to $i = d$) have d elements and improvement is not possible. Note that Caltech hypercube has dimension 5 and Intel hypercube has dimension 7, [7].

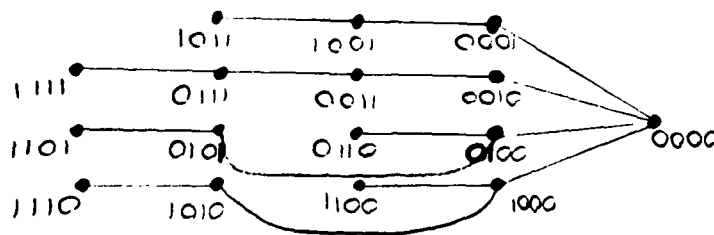


Figure 19: Improvement of Multinode Broadcast Tree for Hypercube $d = 4$

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